Yealink
easy VoIP

SIP-T48G

Revolutionary SIP phone with a 7-inch touch-panel function

The SIP-T48G is Yealink's most recent innovative IP Phone for a fast-changing world. Designed specifically for both local and international use by business, industry and commerce, it incorporates a large touch panel that makes switching between different screens and applications swift, easy and convenient. Yealink's Optima HD technology very carefully creates the impression that you are sitting virtually face-to-face opposite the person at the other end of the call. The T48G is also built for the Gigabit Ethernet and facilitates very rapid call handling and the application of accessories such as a Bluetooth USB Dongle, plus wired and wireless headsets. The new IP Phone supports impressive productivity-enhancing features too which make it the natural and obvious efficiency tool for today's busy executives and professionals.

Optima HD Voice | 7 inch touch screen | Gigabit | Paperless | Support Bluetooth Earphone

Key Features and Benefits

Revolutionarily new design

Yealink 's SIP Phones continue to evolve, the SIP-T4 Series have been designed by pursuing ease of use in even the tiniest details, these new design include 7 inch touch screen, scratch-resistant surface, metallic texture, non-slip rubber feet, ergonomic recessed buttons etc.

HD Audio

Yealink Optima HD Voice refers to the combination of software and hardware design as well as the implementation of wideband technology to maximizes the acoustic performance. Coupled with advanced acoustic clarity technology such as full duplex, echo cancellation, adaptive jitter buffer etc. Creating an amazing face-to-face live experience.

A rich visual experience for applications

SIP-T48G Supports vast productivity-enhancing feature such as XML Browser, call forward, call transfer, 3-way conferencing and a superior UC experience. the 7" 800 x 480-pixel color touch screen with backlight enables rich visual presentation and easier navigation of the menu.

Advanced connectivity

For network connectivity, the SIP-T48G comes with two Gigabit Ethernet ports, one of them suitable for Power over Ethernet. Built-in a USB port, support Bluetooth headset through USB Dongle.

Highly secure transport and interoperability

The Communicator uses SIP over Transport Layer Security (TLS/SSL) to provide service providers the latest technology for enhanced network security. The range is certified compatible with 3CX, Asterisk and BroadSoft Broadworks, ensuring excellent compatibility with leading soft switch suppliers.

- > Revolutionarily new design
- > Yealink Optima HD voice
- > Dual-port Gigabit Ethernet
- > PoE support
- > 7" 800 x 480-pixel color touch screen with backlight
- > Support Bluetooth headset through USB Dongle
- > Up to 6 SIP accounts
- > Paper label free design
- > Headset, EHS support
- > Supports expansion modules
- > Wall mountable
- > Simple, flexible and secure provisioning options

SIP-T48G Specifications

Audio Features

> HD voice: HD handset, HD speaker
> Codecs: G.722, G.711(A/μ), G.723.1, G.729AB, G.726, GSM, iLBC
> DTMF: In-band, Out-of-band(RFC 2833) and SIP INFO
> Full-duplex hands-free speakerphone with AEC
> VAD, CNG, AEC, PLC, AJB, AGC

Phone Features

> 6 VoIP accounts
> Call hold, mute, DND
> One-touch speed dial, hotline
> Call forward, call waiting, call transfer
> Group listening, emergency call
> Redial, call return, auto answer
> 3-way conferencing
> Direct IP call without SIP proxy
> Ring tone selection/import/delete
> Set date time manually or automatically
> Dial plan, XML Browser, action URL/URI

Directory

> Loal phonebook up to 1000 entries
> Black list
> XML/LDAP remote phonebook
> Intelligent search method
> Phonebook search/import/export
> Call history: dialed/received/missed/forwarded

IP-PBX Features

> Busy Lamp Field (BLF)
> Bridged Line Apperance(BLA)
> Anonymous call, anonymous call rejection
> Remote office, hot-desking
> Message Waiting Indicator (MWI)
> Voice mail, call park, call pickup
> Intercom, paging, music on hold
> Call completion, call recording

Display and Indicator

> 7" 800 x 480-pixel color touch screen with backlight
> 24 bit depth color
> LED for call and message waiting indication
> Wallpaper
> Intuitive user interface with icons and soft keys
> National language selection
> Caller ID with name, number and photo

Feature keys

> 29 one-touch DSS keys
> 7 features keys: message, headset, hold, mute, transfer, redial, hands-free speakerphone
> 6 navigation keys
> Volume control keys
> Illuminated mute key
> Illuminated headset key
> Illuminated hands-free speakerphone key

Interface

> Dual-port Gigabit Ethernet
> Power over Ethernet (IEEE 802.3af), class 3
> Built-in a USB port, support Bluetooth headset (Through USB Dongle)
> 1xRJ9 (4P4C) handset port
> 1xRJ9 (4P4C) headset port
> 1XRJ12 (6P6C) EXT port:
Supports up to 6 Expansion Modules for an attendant console application

Other Physical Features

> Wall mountable (optional)
> External universal AC adapter (optional):
AC 100~240V input and DC 5V/2A output
> Power consumption (PSU): 2.0-6.4W
> Power consumption (PoE): 2.4-10.5W
> Dimension(W*D*H*T):
266mm*226mm*185mm*54mm
> Operating humidity: 10~95%
> Storage temperature: -10~50°C

Management

> Configuration: browser/phone/auto-provision
> Auto provision via FTP/TFTP/HTTP/HTTPS for mass deploy
> Auto-provision with PnP
> Zero-sp-touch, TR-069
> Phone lock for personal privacy protection
> Reset to factory, reboot
> Package tracing export, system log

Network and Security

> SIP v1 (RFC2543), v2 (RFC3261)
> Call server redundancy supported
> NAT transverse: STUN mode
> Proxy mode and peer-to-peer SIP link mode
> IP assignment: static/DHCP/PPPoE
> HTTP/HTTPS web server
> Time and date synchronization using SNTP
> UDP/TCP/DNS-SRV(RFC 3263)
> QoS: 802.1p/Q tagging (VLAN), Layer 3 ToS DSCP
> SRTP for voice
> Transport Layer Security (TLS)
> HTTPS certificate manager
> AES encryption for configuration file
> Digest authentication using MD5/MD5-sess
> OpenVPN, IEEE802.1X
> IPv6

Package Features

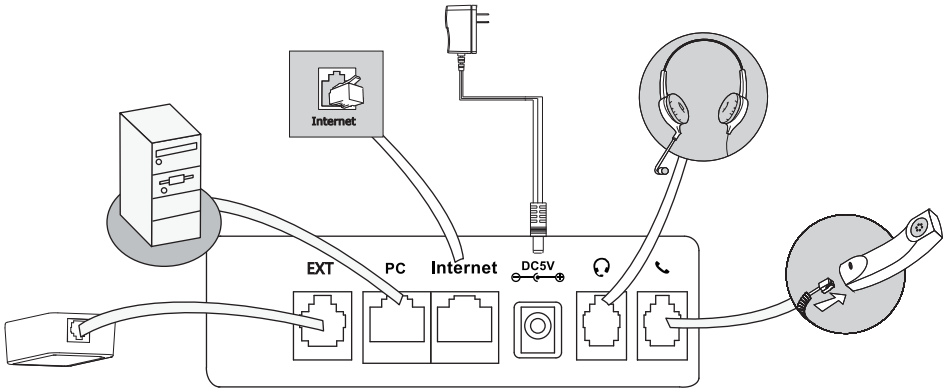
> Qty/CTN: 5 PCS
> N.W/CTN:
> G.W/CTN:
> Giftbox size: 324mm*263mm*128mm
> Carton Meas: 660mm*338mm*273mm

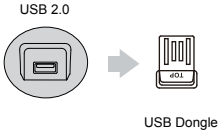
Certifications

CE FC RoHS REACH ISO 9001

3CX Asterisk AVAYA BROADSOFT

Genesys





Learn More

To find out how Yealink solutions can help your organization, visit us at www.e-netsource.com or mail to sales@e-netsource.com

Yealink
easy VoIP

SIP-T46G

Revolutionary SIP phone with a multitude of professional features

The SIP-T46G is Yealink latest revolutionary IP Phone for executive users and busy professionals. New designs appears the commerce, with high-resolution TFT color display, delivering a rich visual experience. Yealink Optima HD technology enables rich, clear, life-like voice communications. Supports Gigabit Ethernet, a variety of device connections, including EHS headset and USB. With programmable keys, the IP Phone supports vast productivity-enhancing features.

Optima HD Voice

Gigabit

Paperless

Support Bluetooth Earphone

Key Features and Benefits

Revolutionarily new design

Yealink 's SIP Phones continue to evolve, the SIP-T4 Series have been designed by pursuing ease of use in even the tiniest details, these new design include paper label free design, new foot stand allows two posi-tions for the device, non-slip rubber feet, ergonomic recessed buttons etc.

HD Audio

Yealink Optima HD Voice refers to the combination of software and hardware design as well as the implementation of wideband technology to maximizes the acoustic performance. Coupled with advanced acoustic clarity technology such as full duplex, echo cancellation, adaptive jitter buffer etc. Creating an amazing face-to-face live experience.

A rich visual experience for applications

SIP-T46G Supports vast productivity-enhancing feature such as XML Browser, SCA, BLF List, call forward, call transfer, 3-way conferencing, the 4.3 inch backlit color display enables rich visual presentation and easier navigation of the menu. Three pages of 10 flexible buttons are shown on the display can be programmed up to 27 various features.

Advanced connectivity and expandability

For network connectivity, the SIP-T46G comes with two Gigabit Ethernet ports, one of them suitable for Power over Ethernet. Supports a variety device connections, including EHS headset and USB. User enable Bluetooth headset through USB Dongle.

Supports up to 6 Expansion Module, adding up to 240 additional buttons with a screen based LCD display and LED system.

Highly secure transport and interoperability

The Communicator uses SIP over Transport Layer Security (TLS/SSL) to provide service providers the latest technology for enhanced network security. The range is certified compatible with 3CX, Asterisk and BroadSoft Broadworks, ensuring excellent compatibility with leading soft switch suppliers.

- > Revolutionarily new design
- > Yealink Optima HD voice
- > Dual-port Gigabit Ethernet
- > 4.3" 480 x 272-pixel color display with backlight
- > Built-in a USB port, support Bluetooth headset (Through USB Dongle)
- > Up to 6 SIP accounts
- > Paper label free design
- > PoE support
- > Headset, EHS support
- > Supports expansion modules
- > Stand with 2 adjustable angles
- > Wall mountable
- > Simple, flexible and secure provisioning options

SIP-T46G Specifications

Audio Features

> HD voice: HD handset, HD speaker
> Codecs: G.722, G.711(A/μ), G.723, G.729AB, G.726, GSM, iLBC
> DTMF: In-band, Out-of-band(RFC 2833) and SIP INFO
> Full-duplex hands-free speakerphone with AEC
> VAD, CNG, AEC, PLC, AJB, AGC

Phone Features

> 6 VoIP accounts
> Call hold, mute, DND
> One-touch speed dial, hotline
> Call forward, call waiting, call transfer
> Group listening, SMS, emergency call
> Redial, call return, auto answer
> 3-way conferencing
> Direct IP call without SIP proxy
> Ring tone selection/import/delete
> Set date time manually or automatically
> Dial plan
> XML Browser
> Action URL/URI

Directory

> Loal phonebook up to 1000 entries
> Black list
> XML/LDAP remote phonebook
> Intelligent search method
> Phonebook search/import/export
> Call history: dialed/received/missed/forwarded

IP-PBX Features

> Busy Lamp Field (BLF)
> Bridged Line Apperance(BLA)
> Anonymous call, anonymous call rejection
> Hot-desking
> Message Waiting Indicator (MWI)
> Voice mail
> Call park, call pickup
> Intercom, paging
> Music on hold
> Call completion
> Call recording

Display and Indicator

> 4.3" 480 x 272-pixel color display with backlight
> 16 bit depth color
> LED for call and message waiting indication
> Dual-color (red or green) illuminated LEDs for line status information
> Wallpaper
> Intuitive user interface with icons and soft keys
> National language selection
> Caller ID with name, number and photo

Feature keys

> 10 line keys with LED
> 10 line keys can be programmed up to 27 various features (3-page view)
> 7 features keys: message, headset, hold, mute, transfer, redial, hands-free speakerphone
> 4 context-sensitive “soft” keys
> 6 navigation keys
> Volume control keys
> Illuminated mute key
> Illuminated headset key
> Illuminated hands-free speakerphone key

Interface

> Dual-port Gigabit Ethernet
> Support Bluetooth headset through USB Dongle
> 1xRJ9 (4P4C) handset port
> 1xRJ9 (4P4C) headset port
> 1XRJ12 (6P6C) EXT port:
Supports up to 6 Expansion Modules for an attendant console application
> Power over Ethernet (IEEE 802.3af), class 3

Other Physical Features

> Stand with 2 adjustable angles
> Wall mountable
> External universal AC adapter (optional):
AC 100~240V input and DC 5V/2A output
> Power consumption (PSU): 1.8-5.4W
> Power consumption (PoE): 2.1-8.0W
> Dimension(W*D*H*T):
244mm*213mm*185mm*54mm
> Operating humidity: 10~95%
> Storage temperature: -10~50°C

Management

> Configuration: browser/phone/auto-provision
> Auto provision via FTP/TFTP/HTTP/HTTPS for mass deploy
> Auto-provision with PnP
> BroadSoft device management
> Zero-sp-touch, TR-069
> Phone lock for personal privacy protection
> Reset to factory, reboot
> Package tracing export, system log

Network and Security

> SIP v1 (RFC2543), v2 (RFC3261)
> Call server redundancy supported
> NAT transverse: STUN mode
> Proxy mode and peer-to-peer SIP link mode
> IP assignment: static/DHCP/PPPoE
> HTTP/HTTPS web server
> Time and date synchronization using SNTP
> UDP/TCP/DNS-SRV(RFC 3263)
> QoS: 802.1p/Q tagging (VLAN), Layer 3 ToS DSCP
> SRTP for voice
> Transport Layer Security (TLS)
> HTTPS certificate manager
> AES encryption for configuration file
> Digest authentication using MD5/MD5-sess
> OpenVPN, IEEE802.1X
> IPv6

Package Features

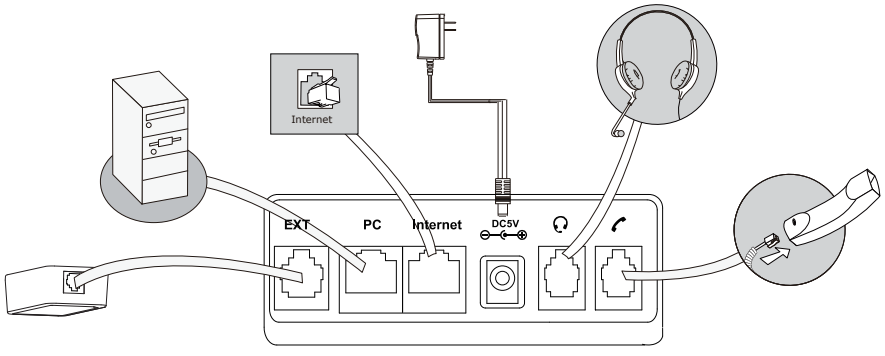
> Qty/CTN: 5 PCS
> N.W/CTN: 8.1KG
> G.W/CTN: 9.16K
> Giftbox size: 274mm*258mm*142mm
> Carton Meas: 732mm*286mm*266mm

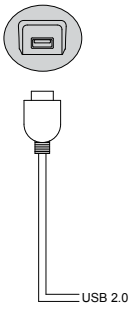
Certifications

REACH ISO 9001

3CX Asterisk AVAYA BROADSOFT

Genesys





Learn More

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Yealink
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SIP-T42G

Standard and Affordable SIP Phone for Business

The SIP-T42G is a feature-rich sip phone for business. The 3-Line IP Phone has been designed by pursuing ease of use in even the tiniest details. Delivering a superb sound quality as well as rich visual experience. Supports seamless migration to GigE-based network infrastructure. With programmable Keys, the IP Phone supports vast productivity-enhancing features. Using standard encryption protocols to perform highly secure remote provisioning and software upgrades.

HD Optima HD Voice

1000M Gigabit

Paperless

Key Features and Benefits

Revolutionarily new design

Yealink's SIP Phones continue to evolve, the T4 Series have been designed by pursuing ease of use in even the tiniest details, these new design include paper label free design, new foot stand allows two posi-tions for the device, non-slip rubber feet, ergonomic recessed buttons etc.

HD Audio

Yealink Optima HD Voice refers to the combination of software and hardware design as well as the implementation of wideband technology to maximizes the acoustic performance. Coupled with advanced acoustic clarity technology such as full duplex, echo cancellation, Adaptive jitter buffer etc. Creating an amazing face-to-face live experience.

Enhanced Call Management

The SIP-T42G supports vast productivity-enhancing feature such as SCA, BLF List, call forward, call transfer, 3-way conference.Three pages of 6 flexible buttons are shown on the display can be programmed up to 15 various features. Support Yealink YHS32, With EHS36 user can control phone through wireless headset.

Efficient Installation and Provisioning

Dual-port Gigabit Ethernet is designed for flexible deployment options and lower cabling expenses. Integrated IEEE 802.3af Power-over-Ethernet allows easy deployment with centralized powering and backup.The SIP-T42G support the FTP, TFTP, HTTP, and HTTPS protocols for file provisioning and are configured by default to use Trivial File Transfer Protocol (TFTP), supports AES encrypted XML configuration file.

Highly secure transport and interoperability

The Communicator uses SIP over Transport Layer Security (TLS/SSL) to provide service providers the latest technology for enhanced network security. The range is certified compatible with 3CX, Asterisk and Broadsoft Broadworks, ensuring excellent compatibility with leading soft switch suppliers.

- > Revolutionarily new design
- > Yealink Optima HD voice
- > Dual-port Gigabit Ethernet
- > 2.7" 192x64-pixel graphical LCD with backlight
- > Up to 3 SIP accounts
- > Paper label free design
- > PoE support
- > Headset, EHS support
- > Integrated stand with 2 adjustable angles
- > Wall mountable
- > Simple, flexible and secure provisioning options

SIP-T42G Specifications

Audio Features

> HD voice: HD handset, HD speaker
> Wideband codec: G.722
> Narrowband codec: G.711(A/μ), G.723.1, G.729AB , G.726, GSM, iLBC
> DTMF: In-band, Out-of-band(RFC 2833) and SIP INFO
> Full-duplex hands-free speakerphone with AEC
> VAD, CNG, AEC, PLC, AJB, AGC

Phone Features

> 3 VoIP accounts
> One-touch speed dial, redial
> Call forward, call waiting
> Call transfer, call hold
> Call return, group listening
> Mute, auto answer, DND
> 3-way conference call
> Direct IP call without SIP proxy
> Ring tone selection/import/delete
> Hotline, emergency call
> Set date time manually or automatically
> Dial plan
> XML Browser,
> Action URL/URI

Directory

> Local phonebook up to 1000 entries
> Black list
> XML/LDAP remote phonebook
> Intelligent search method
> Phonebook search/import/export
> Call history: dialed/received/missed/forwarded

IP-PBX Features

> Busy Lamp Field (BLF)
> Bridged Line Apperance(BLA)
> Anonymous call, anonymous call rejection
> Hot-desking
> Message Waiting Indicator (MWI)
> Voice mail
> Call park, call pickup
> Intercom, paging,
> Music on hold
> Call recording

Display and Indicator

> 2.7" 192x64-pixel graphical LCD with backlight
> LED for call and message waiting indication
> Dual-color (red or green) illuminated LEDs for line status information
> Intuitive user interface with icons and soft keys
> National language selection
> Caller ID with name, number

Feature keys

> 6 line keys with LED
> 6 line keys can be programmed up to 15 various features (3-page view)
> 5 features keys: message, headset, mute, redial, hands-free speakerphone
> 4 context-sensitive “soft” keys
> 6 navigation keys
> 2 volume control keys
> Illuminated mute key
> Illuminated headset key
> Illuminated hands-free speakerphone key

Interface

> Dual-port Gigabit Ethernet
> 1xRJ9 (4P4C) handset port
> 1xRJ9 (4P4C) headset port
> 1XRJ12 (6P6C) EHS port
> Power over Ethernet (IEEE 802.3af), Class 2

Other Physical Features

> Stand with 2 adjustable angles
> Wall mountable
> External universal AC adapter (optional): AC 100~240V input and DC 5V/1.2A output
> Power consumption (PSU): 1.4W-3.9W
> Dimension(W*D*H*T): 212mm*189mm*175mm*54mm
> Operating humidity: 10~95%
> Storage temperature: -10~50°C

Package Features

> Qty/CTN: 5 PCS
> N.W/CTN: 6.9 kg
> G.W/CTN: 7.6 kg
> Giftbox size: 241mm*236mm*138mm
> Carton Meas: 707mm*253mm*244mm

Management

> Configuration: browser/phone/auto-provision
> Auto provision via FTP/TFTP/HTTP/HTTPS for mass deploy
> Auto-provision with PnP
> Zero-sp-touch
> Phone lock for personal privacy protection
> Reset to factory, reboot
> Package tracing export, system log

Network and Security

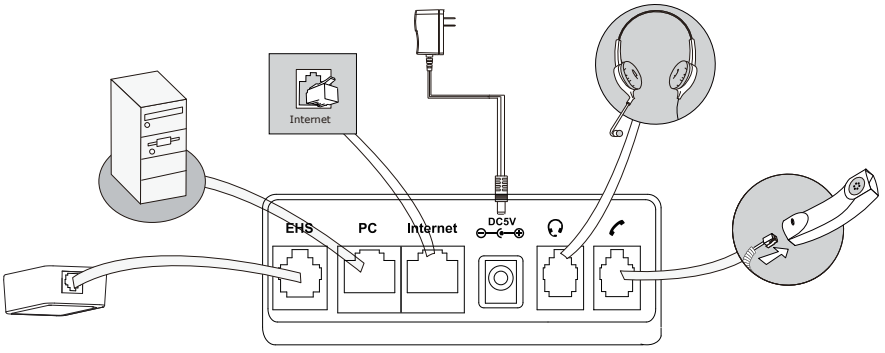
> SIP v1 (RFC2543), v2 (RFC3261)
> Call server redundancy supported
> NAT transverse: STUN mode
> Proxy mode and peer-to-peer SIP link mode
> IP assignment: static/DHCP
> HTTP/HTTPS web server
> Time and date synchronization using SNTP
> UDP/TCP/DNS-SRV(RFC 3263)
> QoS: 802.1p/Q tagging (VLAN), Layer 3 ToS DSCP
> SRTP for voice
> Transport Layer Security (TLS)
> HTTPS certificate manager
> AES encryption for configuration file
> Digest authentication using MD5/MD5-sess
> OpenVPN, IEEE802.1X
> IPv6

Certifications

CE FC RoHS REACH ISO 9001

3CX Asterisk AVAYA BROADSOFT

Genesys



Learn More

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Yealink
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SIP-T41P

Affordable SIP Phone for clear communications

The SIP-T41P is a feature-rich sip phone for business. The 3-Line IP Phone has been designed by pursuing ease of use in even the tiniest details. Delivering a superb sound quality as well as rich visual experience. With programmable Keys, the IP Phone supports vast productivity-enhancing features. Using standard encryption protocols to perform highly secure remote provisioning and software upgrades.

HD Optima HD Voice

Paperless

Key Features and Benefits

Revolutionarily new design

Yealink's SIP Phones continue to evolve, the T4 Series have been designed by pursuing ease of use in even the tiniest details, these new design include paper label free design, new foot stand allows two posi-tions for the device, non-slip rubber feet, ergonomic recessed buttons etc.

HD Audio

Yealink Optima HD Voice refers to the combination of software and hardware design as well as the implementation of wideband technology to maximizes the acoustic performance. Coupled with advanced acoustic clarity technology such as full duplex, echo cancellation, Adaptive jitter buffer etc. Creating an amazing face-to-face live experience.

Enhanced Call Management

The SIP-T41P supports vast productivity-enhancing feature such as SCA, BLF List, call forward, call transfer, 3-way conference.Three pages of 6 flexible buttons are shown on the display can be programmed up to 15 various features. Support Yealink YHS32, With EHS36 user can control phone through wireless headset.

Efficient Installation and Provisioning

Integrated IEEE 802.3af Power-over-Ethernet allows easy deployment with centralized powering and backup.The SIP-T41P support the FTP, TFTP, HTTP, and HTTPS protocols for file provisioning and are configured by default to use Trivial File Transfer Protocol (TFTP), supports AES encrypted XML configuration file.

Highly secure transport and interoperability

The Communicator uses SIP over Transport Layer Security (TLS/SSL) to provide service providers the latest technology for enhanced network security. The range is certified compatible with 3CX, Asterisk and BroadSoft Broadworks, ensuring excellent compatibility with leading soft switch suppliers.

- > Revolutionarily new design
- > Yealink Optima HD voice
- > 2.7" 192x64-pixel graphical LCD with backlight
- > Up to 3 SIP accounts
- > Paper label free design
- > PoE support
- > Headset, EHS support
- > Integrated stand with 2 adjustable angles
- > Wall mountable
- > Simple, flexible and secure provisioning options

SIP-T41P Specifications

Audio Features

> HD voice: HD handset, HD speaker
> Wideband codec: G.722
> Narrowband codec: G.711(A/μ), G.723.1, G.729AB , G.726, GSM
> DTMF: In-band, Out-of-band(RFC 2833) and SIP INFO
> Full-duplex hands-free speakerphone with AEC
> VAD, CNG, AEC, PLC, AJB, AGC

Phone Features

> 3 VoIP accounts
> One-touch speed dial, redial
> Call forward, call waiting
> Call transfer, call hold
> Call return, group listening
> Mute, auto answer, DND
> 3-way conference call
> Direct IP call without SIP proxy
> Ring tone selection/import/delete
> Hotline, emergency call
> Set date time manually or automatically
> Dial Plan
> XML Browser
> Action URL/URI

Directory

> Local phonebook up to 1000 entries
> Black list
> XML/LDAP remote phonebook
> Intelligent search method
> Phonebook search/import/export
> Call history: dialed/received/missed/forwarded

IP-PBX and BroadSoft Features

> Busy Lamp Field (BLF)
> Bridged Line Apperance(BLA)
> Anonymous call, anonymous call rejection
> Hot-desking
> Message Waiting Indicator (MWI)
> Voice mail
> Call park, call pickup
> Intercom, paging
> Music on hold
> Call recording

Display and Indicator

> 2.7" 192x64-pixel graphical LCD with backlight
> LED for call and message waiting indication
> Dual-color (red or green) illuminated LEDs for line status information
> Intuitive user interface with icons and soft keys
> National language selection
> Caller ID with name, number

Feature keys

> 6 line keys with LED
> 6 line keys can be programmed up to 15 various features (3-page view)
> 5 features keys: message, headset, mute, redial, hands-free speakerphone
> 4 context-sensitive “soft” keys
> 6 navigation keys
> 2 volume control keys
> Illuminated mute key
> Illuminated headset key
> Illuminated hands-free speakerphone key

Interface

> 2xRJ45 10/100M Ethernet ports
> 1xRJ9 (4P4C) handset port
> 1xRJ9 (4P4C) headset port
> 1XRJ12 (6P6C) EHS port
> Power over Ethernet (IEEE 802.3af), Class 2

Other Physical Features

> Stand with 2 adjustable angles
> Wall mountable
> External universal AC adapter (optional): AC 100~240V input and DC 5V/1.2A output
> Power consumption (PSU): 1.13W-2.31W
> Dimension(W*D*H*T): 212mm*189mm*175mm*54mm
> Operating humidity: 10~95%
> Storage temperature: -10~50°C

Package Features

> Qty/CTN: 5 PCS
> N.W/CTN: 6.9 kg
> G.W/CTN: 7.6 kg
> Giftbox size: 241mm*236mm*138mm
> Carton Meas: 707mm*253mm*244mm

Management

> Configuration: browser/phone/auto-provision
> Auto provision via FTP/TFTP/HTTP/HTTPS for mass deploy
> Auto-provision with PnP
> BroadSoft device management
> Zero-sp-touch
> Phone lock for personal privacy protection
> Reset to factory, reboot
> Package tracing export, system log

Network and Security

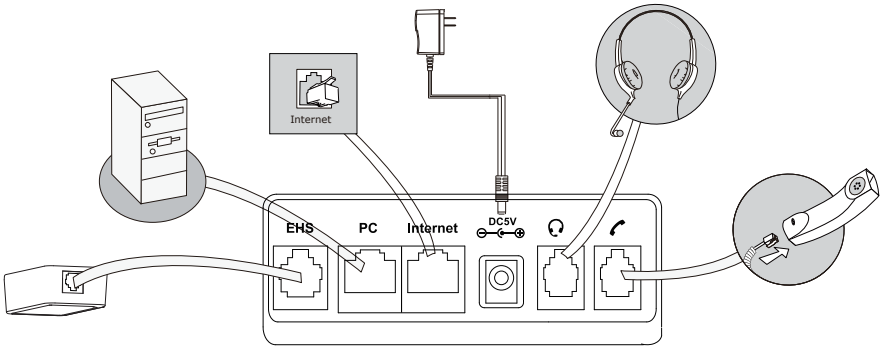
> SIP v1 (RFC2543), v2 (RFC3261)
> Call server redundancy supported
> NAT transverse: STUN mode
> Proxy mode and peer-to-peer SIP link mode
> IP assignment: static/DHCP
> HTTP/HTTPS web server
> Time and date synchronization using SNTP
> UDP/TCP/DNS-SRV(RFC 3263)
> QoS: 802.1p/Q tagging (VLAN), Layer 3 ToS DSCP
> SRTP for voice
> Transport Layer Security (TLS)
> HTTPS certificate manager
> AES encryption for configuration file
> Digest authentication using MD5/MD5-sess
> OpenVPN, IEEE802.1X
> IPv6

Certifications

REACH ISO 9001

3CX Asterisk AVAYA BROADSOFT

Genesys



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